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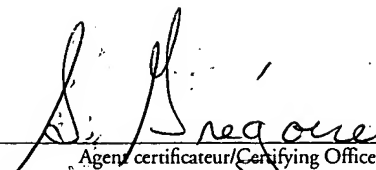
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Specification and Drawings, as originally filed, with Application for Patent Serial No:
2,324,572, on October 26, 2000, by GERRY M. KANE, for "Digital Vibration
Transducer".


Agent certificateur/Certifying Officer
November 20, 2001

Date

Canada

(CIPO 68)
01-12-00

OPIC  CIPO

DIGITAL VIBRATION TRANSDUCER

SCOPE OF THE INVENTION

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The invention is a device that, by using the principle of optical leveraging and an optical digital encoding sensor, detects vibrational phenomena from such sources as audio, seismic, hydrophonic, barometric or other cyclic motion sources and directly converts that motion into a digital signal for the purpose of recording, amplification, analysis, processing, display or entertainment. Since the transducer is optical in nature, there is no analog electronic stage, no coils, magnets or condensers and thus an extremely high quality of performance becomes possible.

BACKGROUND OF THE INVENTION

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The existing art of analog transducers, such as speakers and microphones, suffers from a number of limitations. Among these limitations are dynamic range, distortion and, for microphones especially, noise.

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Dynamic range is the range of volume that the transducer can detect or, in the case of a speaker, produce. But it is also important to distinguish between overall dynamic range and instantaneous dynamic range. Thus while the human ear can easily have an overall dynamic range of 120 dB and can, in a quiet room, hear a pin drop and at another time accurately hear the notes playing at a 120 dB rock concert, it can't hear both sounds simultaneously. This is also true of analog microphones. If one raises the gain of a sensitive analog microphone high enough it can pick up the sound of a pin dropping. But one cannot then expect the same microphone, at the same gain, to cleanly detect sound at 120 dB. Conversely, an analog microphone attenuated enough to detect a 120dB concert without clipping (cutting off the peaks of the wave forms) would be too insensitive to detect even

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normal conversation adequately. Thus, like the human ear an analog microphone cannot detect both a very loud sound and a soft sound simultaneously.

The digital vibration transducer (DVT) can be used to construct a microphone with an instantaneous dynamic range of more than 120 dB. The dynamic range of a digital device is determined by the number of bits utilized or quantization level. Thus, for a compact disc, which uses 16 bit quantization, the dynamic range is 96 dB, which corresponds to a 65,535 fold difference in volume. The addition of another bit to make a 17 bit quantization device would double that range to 102 dB or 131,070 fold. At 18 bits quantization the dynamic range is 108 dB or 262,140 fold, at 20 bits the dynamic range is 120 dB or 1,048,560 fold, and at 22 bits the range is 132 dB or 4,194,240 fold. At 24 bits (the present standard for professional mixing) the dynamic range is 144 dB or 16,776,960 fold.

While there is no absolute limit to the maximum quantization level possible for the digital vibration transducer, the ideal level varies with the application. When the vibration source is vocal sound, particularly vocal music, and the application is as a microphone such as a hand held or headset mounted microphone, and a high sensitivity is desired, it is necessary for most parts of the device to be as small and light as practical. A quantization level of 20 bits would appear to be best for constructing a DVT microphone having dimensions somewhat smaller than a C cell battery, light overall weight, and high sensitivity. Such a microphone would have a minimum 120 dB dynamic range and since this would be the instantaneous dynamic range, such a microphone detecting a sound of 120 dB would theoretically also be capable of detecting a sound of 0 dB at the same time. In practice, however, it would be desirable for such a microphone, if intended for loud environments, to have some "headroom" and be able to withstand a sound pressure level (SPL) of 140 dB. Also, as described further below, the first 20 dB above the detection threshold of such a microphone can have distortion over 5 %. It is, therefore, more realistic to say that the usable dynamic range of such a microphone would be, at least, 100 dB and that in a loud environment, if so

adjusted, it would be capable of faithfully capturing sounds in the range of 40 dB to 140 dB simultaneously. Since normal conversation is considered to be at 74 dB, someone speaking in a normal voice a few feet from this microphone, in a loud environment, would be detected, and what was said could be extracted from a digital recording, even though a person standing right next to the one speaking wouldn't have been able to hear a word.

This extraordinary capability has implications beyond those that might be immediately apparent. Such a microphone placed in the cockpit of an aircraft, for example, or used to replace the relatively limited quality microphones on the flight officers' headsets, would allow for cockpit voice recordings that include much greater detailed and accurate sounds of the environment, and which would also be more easily processed to extract isolated sounds.

When size is of less concern the DVT can easily be constructed to provide a 24-bit quantization level. This with a sensitivity well below 0 dB and while retaining a size not larger than two D cell batteries. Such a DVT in the application of a hydrophone would be able to detect weak underwater sounds even when the props and engines of a nearby ship are also being detected and would swamp a conventional analog hydrophone.

In seismology the mass of the vibration source (the earth) is so great that the mass and size of the components in the DVT becomes of lesser concern and quantization levels over 24 bits are easily achieved. A 24-bit quantization level corresponds to a 144 dB instantaneous dynamic range or 16,777,960 times, which is also equivalent to about 8 points of magnitude on the Richter scale. Thus, the DVT used as a seismograph could be expected to easily and accurately detect vibrations differing by 7 points of magnitude simultaneously. A typical analog seismograph, designed to continuously monitor events at around magnitude 2 is swamped by a magnitude 7 event and will lose detail of any lower magnitude vibrations happening at the same time. It is conceivable that a DVT seismograph

might reveal details of the earth dynamics of which we are, as yet, unaware. Further such a device interfaces well with existing systems that record their data in a digital format, on a hard drive for example, so that continuous high resolution monitoring can be maintained. Such existing systems currently require converting
5 analog signals from analog seismographs by means of an ADC, and have their instantaneous dynamic range thus limited to that available from the analog stage (about 40 dB or 100 times).

Another outstanding feature of the DVT is its frequency response, especially its
10 low frequency response. Thus, while there may be an upper frequency limitation, and this limitation is dependent on the mass of certain components, there is no lower limit and frequencies as low as 0.1 Hz and lower are detectable. Depending on how the sensing membrane is vented to the atmosphere the device can be used as an extremely sensitive barometric device such as a variometer.

15 A second limitation of the existing art of analog sound transducers is distortion and this has a relationship to dynamic range. Generally speaking, analog transducers have more distortion the higher they are in their dynamic range. Thus the louder the sound entering a microphone or leaving a speaker, for example, the more
20 likely it is to distort. A major source of this distortion is due to non-linearity in the suspension of the sound sensing or sound invoking membrane. In order for a sound sensing or invoking membrane to have a rest position to which it seeks and will settle when there is no signal, the suspension must have an elastic, non-linear nature. That is, the farther the membrane is deviated from its rest position the
25 greater must be the opposing force of the suspension attempting to return it. It is well known that sound passing through a non-linear medium will suffer distortion as a result and if the sound is composed of more than one frequency inter-modulation (IM) distortion results causing the introduction of tones not in the
30 original signal.

While a DVT, when used for such applications, will also be prone to the same distortions inherent in devices using suspended membranes, the distortions can be removed by calibrative compensation. The DVT attached to a sound sensing membrane, for example, will output a number that corresponds to the exact position of the membrane. Since the non linearity of the suspension is predictable it is possible to calculate how far off the position of the membrane is at any time as compared to where it would be were the suspension perfectly linear. Thus, by adjusting the digital vibration detector's output with a simple processor using a calibrating algorithm, or by using a PROM (programmable read only memory) to act as a calibration "chart", virtually all distortion can be removed in the result.

The distortion that cannot be removed in this way is the distortion of quantization. When an analog signal is digitized it is chopped up into discrete levels that approximate the original signal. The accuracy of that approximation depends on the level of quantization, how many different levels the signal can be chopped into. Thus, if a signal is so weak it only just exceeds the threshold of detection, the level of quantization for that signal may be as low as one bit, one level, either on or off. The output of the device in that case can only be a square wave. If the input signal is also a square wave then there may be no distortion but if the input signal is a sine wave then we could say the distortion in that case is perhaps less than 50 %. As the input level of the signal increases, however, more bits come into play and once the signal is strong enough to activate the first 4 bits there will be 15 levels into which it can be chopped and the distortion will have dropped to around 1/15 or less than 3.3 %. Thus, in contrast to analog devices, the higher or louder the input, the lower the distortion, provided the upper limit of volume is not exceeded. Each 20 dB of SPL represents a 10 fold increase of digitization, so for a signal of 20 dB above threshold, distortion is less than 5 %, at 40 dB above threshold - less than 0.5 %, and at 60 dB above threshold, distortion will be less 0.05 %. A 22 bit DVT used as a microphone and having a dynamic range of 132 dB will, therefore, have the 92 dB of that range at a distortion of less than 0.5 % and 72 dB of that range at

less than 0.05 % distortion. This far out performs conventional, analog, microphones only the very best of which can achieve a distortion as low as 1 %.

5 The results are just as dramatic when the DVT is used in attachment with a speaker. In this case the DVT is attached to the speaker cone and used to measure the position of the cone in comparison to where it should be based on the signal being input to the speaker's voice coil. If there is a discrepancy between the output of the DVT and the input signal to the voice coil, a simple comparator circuit develops a correction signal, which is used to adjust the voltage on the speaker's
10 voice coil and thus correct the errors and distortion in the cone's movement whatever their origins may be. This can be achieved with relatively simple electronics added within the speaker enclosure, and without an additional power source if the system uses and stores some of the power already being supplied to the speaker by the power amplifier. This effectively constitutes the digitization of a
15 speaker, and the result of retrofitting even the best existing speaker models with such a system can be expected to exceed a tenfold reduction in distortion.

In regards to noise, the advantage of the DVT over that of its analog counterpart is particularly notable in its application as a microphone or hydrophone. The signal to
20 noise ratio of a digital device is generally the same as its dynamic range or, in the case of a 20-bit device the s/n ratio will be 120 dB. An 80 dB s/n ratio would be considered outstanding for an analog microphone. In the DVT there is no stage where the signal exists as an analog electrical signal so there is no opportunity for the introduction of electrical noise. Similarly, since there are no electrical coils or
25 analog lines of length the device is essentially immune to magnetic fields and all but the most severe electromagnetic interference. This, again, far exceeds the performance of existing analog microphones, for example.

It should also be noted that even though the DVT can be used to construct audio
30 transducers having a many fold improvement in performance over prior technology, that does not necessarily also mean a many fold increase in cost. The

cost of a microphone constructed with the DVT would be comparable to the best examples of its analog counterpart. Retrofitting a speaker for digital correction by the device could be done at percentage of the initial cost of the speaker and could, further, enable less expensive speaker to have a performance exceeding those in a higher price range.

SUMMARY OF THE INVENTION

The primary objective of the present invention is to take advantage of an optical means to translate a mechanical motion into a digital signal without an intervening analog electronic stage. That is, the digital encoding of the motion takes place while the signal is optical in nature and thus the limitations of analog electronics can be avoided. This is achieved by utilizing a mirror mounted on bearings so that the motion to be detected can be linked to the mirror in such a way as to cause it to rotate. The rotation of this mirror optically levers a focused laser source incident upon it causing that laser to sweep across a digital encoding light motion sensor positioned at the focus point of the laser. The digitally encoding sensor operates by translating the movement of the levered light into electrical pulses, the number of which corresponds to the amount of movement of the levered light. The digital encoding sensor is composed of a digital encoding plate and a light sensor placed behind the encoding plate. The encoding plate can be a strip of photographic film that has been exposed to form regularly spaced opaque vertical stripes (throughout this document vertical or height denotes the orientation of the mirror's axis of rotation, and horizontal or width denotes the orientation along which the levered light sweeps). The encoding plate thus has the appearance of a long "picket fence" and as the focused laser spot sweeps horizontally across these vertical pickets the light is alternately blocked and then permitted to pass through the encoding plate. This causes the generation of an electrical pulse, from the light sensor behind the encoding plate, each time the light is able to pass through the encoding plate. It is then a straightforward matter for a binary logic counter to

count these pulses and continuously output a binary number that is a direct representation of the angular position of the rotatable mirror and, thus, the position of the motion source attached to the mirror.

5 This device also incorporates opto-mechanical amplification of the motion source being detected. The amount of this gain is determined by the relationship between the lengths of the levered optical arm and the lever arm represented by the distance from the mirror's axis to the point on the mirror at which the detected motion is being applied.

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The resolution or sensitivity of the device is determined by a number of interdependent parameters. These are:

15 encoder pitch,
laser spot size at the focus,
mirror size (width),
mirror inertia,
opto-mechanical gain.

20 Thus, for the highest resolution it is desired to have the smallest possible pitch (the distance between adjacent pickets and also the width of a picket itself) on the encoding plate. Since this pitch must closely match the focused spot size of the laser, the pitch is determined by the smallest possible spot size to which the laser can be focused. This is further dependent on the width of the rotatable mirror, a
25 greater mirror width being necessary for a smaller spot size (down to the theoretical limit for the particular wavelength of light involved). The wider the mirror, however, the greater the rotational inertia may become, and this reduces sensitivity. Thus, in addition to the basic embodiment, several approaches are described to optimize these interdependencies in regards to the particular
30 application for which the invention is to be used.

In one approach several, independent, digital encoding sensors are used, vertically adjacent to each other and slightly offset from each other horizontally. The laser spot is then optically elongated to form a narrow vertical line so that all of the encoding plates are illuminated. This has the result of decreasing the effective pitch and, consequently, increasing the resolution and sensitivity.

In another approach the rotatable mirror is constructed from more than one reflective piece attached to a lightweight framework. This allows for a lower rotational inertia than would be possible with a single, wide, mirror, and thus, a greater sensitivity can be achieved.

Though it will not be discussed further on, it may also be possible, when the size of the device is not a prime concern, to use a gas type ultraviolet laser rather than the solid-state diode laser presumed. The shorter wavelength of an ultraviolet laser would allow for a smaller dot size and the consequent increase of resolution.

It is when the vibration source is sound and the application of the device is as a microphone or hydrophone that the greatest sensitivity is desired. The larger the size of the device can be, the easier it is to have a high sensitivity. If the application is as a hand held, or headset mounted, microphone it is desirable to make the device as small as practical and that would tend to limit the sensitivity. However, a hand held, or headset mounted, microphone does not need to be as sensitive as microphones used in other applications because the sound source (singer, announcer etc.) will always be relatively close to the microphone, and indeed, it's more desirable for such a microphone to be able to operate well under high sound pressure levels. The description of the embodiments and the technical discussion gives an overview on addressing these types of application requirements.

BRIEF DESCRIPTION OF THE DRAWINGS

Figure 1 is a three dimensional side view of the first embodiment of the device showing a central light ray from the laser light source, and through the optics.

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Figure 2 is a top view of the first embodiment of the device showing the side most light rays from the laser light source and through the optics.

Figure 3A is a face on view of the encoding plate as described in the first embodiment of the device.

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Figure 3B is a face on view of a three level «stacked» encoding plate as described in the second embodiment of the device.

Figure 4A is a portion of the device, from a three-dimensional side view, showing the placement of a cylindrical lens being used to cause a vertical dispersal of the reflected light for possible application in the second embodiment of the device.

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Figure 4B is a portion of the device, from a three-dimensional side view, showing the use of a curved mirror to cause a vertical dispersal of the reflected light for possible application in the second embodiment of the device.

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Figure 5 shows the preferred optical arrangement the device, from a three dimensional side view, for use in the second embodiment of the device, and which show cylindrical lenses being used to constitute the focusing collimator so that the laser light source retains a vertical height upon reaching the encoding plate.

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Figure 6A shows a three level «stacked» light sensor, compatible with the encoding plate in figure 3B, connected to the circuit elements which provide the logic processing of the outputs from the light sensor, and in 6B, a timing chart

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which, further, shows the wave forms to be found at indicated (by italicized letters) points about the circuitry.

Figure 7A shows the top view of the device, as configured with the split mirror variation described in the third embodiment of the device, along with the indicative pathways of a number of the source laser's light rays as they pass through the cylindrical optics preferred in the second embodiment, and three prisms intended to split and redirect the source laser's output beam.

Figure 7B shows a face on view of the split mirrors, in attachment to the frame piece that holds them in position, along with some of the other associated components.

Figure 8 shows the device used in attachment to a speaker, along with certain essential circuit elements, and some sound system components.

DISCRIPTION OF THE PREFERRED EMBODIMENTS

In the fundamental embodiment of the device (figure 1), a first surface mirror (10) is made able to rotate by suspending it between two "vee" bearings (11). These bearings are aligned to permit rotation about a vertical axis horizontally centered on the mirror (throughout these descriptions vertical or height denotes the orientation of the mirror's axis of rotation, and horizontal or width denotes the orientation along which the levered light sweeps). The bearings are presumed to be held in position by a stationary frame (not shown) firmly mounted to the device housing (not shown) which also holds the other, non-dynamic, components in fixed positions. The frame is presumed to cause the bearings to exert a light pressure on the mirror. Since it is desirable to have as little friction and end play as possible (for optimum response and sensitivity), it is preferable to use high precision jeweled bearings. To assist visualization it should be presumed, for figure 1, that

there are tiny holes in the edges of the mirror into which the points of the bearings are seated. For an application such as a microphone, it is further desirable that the mirror has as low a mass as possible without sacrificing stiffness and optical flatness. A thin Pyrex optical glass, vacuum deposited with a reflective metal such as aluminum or gold is one possibility.

A solid-state diode laser (12) operating in its primary mode is directed to this mirror after passing through collimating and focusing optics (13 and 14). The first lens (13) will be called the dispersing lens, the second lens (14), which is taken to be both the final element of the collimator and the focusing objective, will be called the objective, and the system of the two lenses together will be called the focusing collimator. The focal lengths of the optics are such that the laser focuses to a spot some distance after reflecting off of the mirror (10). In figure 1 only the path of the center most light ray (20) of the laser is shown, and this ray also represents the optical axis (20) of the focusing collimator. In figure 2 only the side most light rays (22 and 23) are shown so as to illustrate the refractions of the focusing collimator, and the focal point occurring at the plane of the encoding plate (18). The orientation of the optics is such that the optical axis of the focusing collimator (13 and 14) intersects with the mirrors rotational axis (dot-dash line in fig. 1) at all times. Further, the orientation is such that, when at rest (no vibrational input to the device), the mirror face is perpendicular to the focusing collimator's axis in the horizontal plane, that is, when viewed from above (fig. 2). Thus, when the mirror is at rest, the angle of azimuth between the central (axial) light ray (20, fig. 1) incident upon the mirror and its reflection (21) is zero. The angle of declination between these two rays (20 and 21) is what ever will be necessary to prevent the components of the device from optically or mechanically obstructing each other.

This optical orientation of zero azimuth, when the mirror is at rest, is a distinctive feature of the present invention as compared with the prior art of optically levered lasers, such as in the field of Atomic Force Microscopy (AFM). Such systems of prior art may use an obtuse angle of azimuth, and may even attempt to "skim" the

laser light off of the mirror such that the optical axis of the incident light does not strike the same point on the mirror at all times, and which, thus, invokes a different principle of optical leveraging than that used by the present invention. The focusing of the laser at a point some distance from the mirror, as well as the full illuminating
5 of the mirror, by the present invention, further distinguishes it from prior art in its principles of operation.

Also applied to the mirror are two linkage bearings (Fig. 1, 15) placed at points (tiny holes in the mirror's edges) some distance horizontally offset from the
10 rotational axis. The axis of the linkage bearing has the same vertical orientation as the mirror's rotational axis and the smaller the distance between the two axes, the greater will be the opto-mechanical gain achieved by the device. The linkage bearings are held in place by the linkage-bearing frame (16). The frame is presumed to spring load the bearings so as to maintain a light bearing pressure
15 upon the mirror. Attached to this frame is a linkage arm (17). The linkage arm will be as long as is needed to prevent the mirror from touching the vibration source when the mirror is rotating. The linkage arm can be a thin paper or other lightweight tube so that it will have high stiffness, low mass and little resonance. To the other end of this linkage arm is attached the vibration source, be it a
20 microphone diaphragm, a speaker cone, a seismographic source, hydrophonic source etc. Whatever the vibration source, the attachment is best made so that, when no vibrations are present, the linkage arm (17) is at a right angle to a line drawn between the mirror axis and the linkage bearing axis, and perpendicular to the mirror axis. Further the orientations are such that the vibrational source moves
25 the linkage arm (17) along its longitudinal direction (arrows). Thus, in the case of a microphone diaphragm, sound vibrations are transferred to the linkage arm, via the diaphragm, which causes the mirror to rotate cyclically, which in turn levers (in the manner of a third class lever) the laser light to sweep an arc along the focal plane. The circle of this focal plane arc has the mirror axis as its center. The opto-
30 mechanical gain of this arrangement is approximately the radius of this circle divided by the distance between the mirror axis and the linkage-bearing axis.

Along the arc of the focal plane is placed the digital encoding light motion sensor (Fig. 1, 18 and 19). The encoding sensor is composed of a digital encoding plate (18) and a light sensor (19). The light sensor (19) can be a silicon cell which outputs a current proportional to the light it receives. As described in the summary, the encoding plate (18, and fig. 3A) is composed of opaque vertical stripes or "pickets" (fig. 3A, 24) horizontally equidistant and of the same width as the light transmissive spaces between them. Figure 3A shows a section of the encoding plate as it might appear greatly magnified and viewed face on. In figure 1, to assist visualization, the pickets are shown as dark vertical stripes, having considerable width, on the encoding plate (18). In practice, the pickets would be so narrow (about 1 micron) as to be invisible to the unaided eye. The width of a picket (fig. 3A, 24) is referred to as the "pitch". The pitch is selected to closely match the spot size of the focused laser so that as the mirror (10) levers the laser spot to sweep across the encoding plate (18), the light is alternately blocked and passed through the encoding plate to the light sensor (19) behind. The light sensor (19) consequently outputs square (or squarish) wave pulses in an amount that corresponds to the distance that the laser spot has moved. A binary counter (fig. 6, U20) then counts these pulses and outputs a continually updating binary number that indicates the position of the laser spot in a typical parallel digital format. The resolution to which this movement is measured is, therefore, determined by the encoder's pitch and the laser's focused spot size on which it depends. Thus, the smaller the laser spot size and encoder pitch, the higher the resolution.

The smallest size to which a laser can be focused depends on the wavelength of the laser (12) and the numerical aperture (NA) of the objective lens (14). The spot size cannot be smaller than the wavelength of the laser's monochromatic light and in the interests of cost and overall size of the device it preferable in these embodiments to use a typical solid-state diode laser. These, with present technology, most commonly operate in the red end of the spectrum and, although a reliable solid state blue laser has recently been developed and may prove an effective alternative, the descriptions in this section will assume a red laser of 0.63

microns wavelength. Present technology also routinely manufactures media (such as compact discs, or CDs) having a 1 micron pitch and, so, this will be taken as a realistic pitch for the digital encoding plate. The numerical aperture of an optical system is defined as the product of the refraction index (in this case 1, for air) and the sine of the angle between the optical axis and the outermost light ray contributing to the imaging. Due to diffraction at the lens aperture, however, the laser "spot" does not have a sharp edge and is, instead, a spot brightest at its center, fading towards the edge and having around it annuli of decreasing brightness. If the spot diameter is taken as the half-intensity diameter, it is found that for $NA = 0.36$ and a wavelength of 0.63 microns, the (half-intensity) spot diameter is 1 micron. This corresponds to an objective lens diameter that is 0.77 times the distance of the lens to the focus point, and the relationship is such that the wider the lens is, compared to the focus distance, the smaller the spot will be. Since the length of the optical lever arm is the distance from the mirror to the focus point it is, more importantly, the mirror that will need an NA of 0.36 to produce a spot of 1 micron diameter. The objective lens will be wider than the mirror in order to illuminate it fully. Thus, for a spot size of 1 micron, the mirror width will be 0.77 of its distance from the focus or encoding plate. However, since the mirror will be deviating from its rest position by up to, perhaps, 10 degrees, its apparent width, relative to the focusing collimator's axis, will reduce the greater that deviation is. Thus, the effective width of the mirror at maximum deviation will be equal to the cosine of the deviation angle times the actual width, or 0.985 times actual width for 10 degrees. So, for a mirror that may deviate 10 degrees, its width should actually be about 0.78 times its distance to the encoding plate in order to sustain a 1 micron spot diameter throughout its deviation. It should be further noted, though, that it is not actually necessary for the laser spot to be circular, with a small diameter, and it will suffice if the laser spot is merely narrow in width. Since the orientation of the pickets is vertical, any vertical elongation of the laser at the focus will be irrelevant as long as the elongation is not so much that it causes excessive laser light to be lost by spilling past the encoding sensor altogether. Consequently, it is also found that the height of the mirror is not a contributing factor in causing

the laser to have narrow width at focus and, rather, it is only the width of the mirror that is essential. Even further, it has been found by the experiments of the inventor that the center of the mirror and the central rays of the optics are not essential for producing a laser focus of narrow width, and, instead, only the outer edges of the mirror and the outermost rays (in the horizontal dimension) are needed to provide the desired focus.

If the application of the device is for detection of vibrations having simple waveforms and large power, the above embodiment would be adequate, and the dimensions and mass of the mirror would not be of the utmost concern. If the input waveforms are square waves or sine waves of a single frequency, the circuitry to which the output of the encoding sensor is attached could easily distinguish one direction of spot movement from the other (since the sensor itself, as described above, does not indicate the direction of movement, only the amount). With such waveforms the spot would stop moving before it changed direction and a logic circuit could use the lack of pulses output at that time as a condition for switching the binary counter from up to down, or vice versa. If, however, the application of the device is for detection of complex waveforms (such as sound) whose amplitudes may also be relatively weak the following embodiment is preferred.

In this next embodiment it is necessary for the laser "spot" to be vertically elongated while remaining horizontally narrow. There are basically three effective ways to achieve this elongation which vary in practicality. In the first method (fig. 4a) a cylindrical lens of concave profile (25) is placed between the mirror (410) and the encoding plate (418) and causes the reflected laser light (21) to disperse vertically while remaining unaffected horizontally. It is also best if this lens (25) is curved in an arc that has the mirror's rotational axis as its center.

In the second, more practical, method (fig. 4b) the mirror (410) itself has a vertical curvature and this causes a vertical dispersal of the reflected laser light (21) while leaving it horizontally unaffected.

In the third, and most practical, method (fig. 5) the focusing collimator's optics (13 and 14 of fig. 1) are composed of cylindrical lenses (fig. 5, 513 and 514) which are horizontally curved but have no vertical curvature (in figure 5 the objective, 514, is shown as composed of two plano-convex cylindrical lenses glued together). Thus, focusing occurs only in the horizontal plane and the laser beam (26), being unaffected vertically, arrives at the mirror with its raw height (figure 5 shows only vertical slice of the laser light, 26, that includes the center, top and bottom most rays). Diode lasers typically emit an elliptical beam having dimensions of about 1 by 4 mm. If the diode is oriented to have its shorter beam dimension (1 mm) vertical it will retain that height, after passing through the cylindrical lenses (513 and 514), upon striking the mirror (510), and if the mirror (510) is also about 1 mm high the reflected beam will then strike the encoding plate (518) retaining this 1 mm of height while having been focused horizontally down to a very narrow width. This 1 mm of vertical height will be sufficient for the following, second, embodiment.

In figure 3B a plurality (three, for this example) of independent encoding picket strips (27) (each of 1 micron pitch) are stacked vertically adjacent to each other on the encoding plate. Each of the picket strips is separated from its neighbor by an opaque horizontal stripe (28). This horizontal stripe serves to minimize light that has been modulated by one picket strip (27) from spilling onto the light sensor (fig. 6A, 29) dedicated to an adjacent picket strip. Each picket strip has behind it an independent light sensor (fig. 6A, 29) so that for the three encoding picket strips there are three corresponding light sensors (29), and the arrangement constitutes a vertical stack of three independent encoding sensors. In figure 6 the crosshatch-patterned strips (29), in the light sensor stack, represent the photosensitive sections of the stack, the clear strips (30) are the conductors for the adjacent photosensitive strips, and the black strips (31) are insulative gaps between the three independent sensors. Each picket strip (27) of figure 3B should be visualized as superimposed over the crosshatch strips (29) of figure 6A so as to constitute the vertical stack of encoding sensors. Each picket strip (fig. 3B, 27) is horizontally offset from those to which it is adjacent by an amount equal to $1/n$ times the pitch

($1/n \times 1$ micron, in this case), where n equals the number of picket strips in the stack. Because the laser is now a vertical line, 1 mm by 1 micron, at the focus the full height of the stack is illuminated (the proportions in figure 3B represent that, for this a pitch of 1 micron, each picket would be 2 microns high, and that the combined height of this stack would be less than 9 microns). Thus, as this laser line is levered, by an input signal, horizontally across this stack the individual outputs of each encoding sensor in the stack will be pulses that are $1/2n$ of a cycle out of phase with those sensors to which it is adjacent. After these individual pulse streams are logic gated together (fig. 6A), the resulting single output (G) will have n pulses for each micron of spot movement. This has the result of reducing the effective pitch to $1/n$ th ($1/3$, for this case) of a micron. Also, for a stack of three or more encoding sensors the direction of spot motion can be readily determined by logic discrimination of the order in which the individual sensors are pulsing.

In figure 6A the outputs of each of the light sensors (29) are considered to be negative going when light strikes the sensor. The upward pointing arrows in the figure represent the positive supply voltage which could be 5 volts, for example. The negative going pulses are inverted by comparators U1, U2 and U3 into square wave pulses A, B and C, which are illustrated on the timing chart, figure 6B.

These waveforms are the result of the focused laser line sweeping at a steady speed from left to right across the encoding plate stack illustrated in figure 3B, which is assumed to be superimposed over the light sensor stack in figure 6A. R4 and R5 set the threshold of the comparators and R1, R2 and R3 are pull up resistors that hold the inverting inputs of the comparators positive when no light is striking the associated photosensitive strip. The outputs (A, B and C) of all three comparators are sent to three, three input AND gates (U4, U5 and U6). U4 has no inverting inputs and its output, wave form D, is positive when A, B and C are all positive. U5 has inputs A and B inverted and C non-inverted, so its output, waveform C, is positive when waveforms A and B are negative and C is positive. U6 has inputs A non-inverted and inputs B and C inverted, so its output, waveform F, is positive when A is positive, and B and C are negative. The outputs of U4, U5

and U6 (waveforms *D*, *E* and *F*) are gated together by U7, a three input OR gate, which consequently outputs waveform *G*. This output is a measurement of the amount of movement of the laser line across the encoding plate (fig. 3B) and has a resolution three times that of any single light sensor in the stack. The three
5 waveforms *D*, *E* and *F* are also sent to direction discriminator U10, which outputs a high or a low at output *H* depending on the order in which *D*, *E* and *F* are occurring. The direction signal, *H*, and the waveform to be counted, *G*, are both sent to counter U20 which will count either up or down, depending on the level of *H*, and continuously output a binary number of (for this example) 20 parallel bits.
10 This 20 bit output is then sent to calibrator U21, which adjusts the value of U20's output to compensate for any non linearities, and outputs a corrected result of 21 parallel bits or more, depending on the degree of non linearity.

This next, third, embodiment is recommended for applications requiring the
15 highest sensitivity. Here (fig. 7A and B) the mirror is composed of two mirror pieces (fig. 7A and B, 710). That is, the center section of the mirror is cut out and replaced by a frame piece (fig. 7A and B, 38) of lighter material (such as graphite composite or titanium composite) that hold the two mirror end pieces in the same plane as if they were one continuous mirror. This arrangement also allows these
20 end mirror pieces (710) to be thinner, front to back, since they don't have as much length needing to be sustained optically flat. This arrangement is consequently much lighter than the solid mirror version and is, likewise, that much more sensitive.

25 Since this central section (38) is no longer made of glass, this also makes it easier to inlay a jewel piece (not shown) into the frame piece (38) for the linkage bearings (715) to be mated to, so that both the top and bottom bearings of the linkage (715) are jewel to jewel. Likewise, the rotational axis bearings (711) can be applied to a jewel rod (32) passing through the frame piece (38) so those top and bottom
30 matings are also jewel-to-jewel and thus the (within tolerance) lifetime of all of the bearings will be great indeed.

To make the most effective use of the laser light, in this case, it is best that the laser is concentrated into two beams (33 and 34), of 1 mm height, as they leave the diverging lens (713) so that they strike at the only the edges of the cylindrical objective lens (714). Thus, no light is wasted on the center of that lens (714), since
5 it would be useless anyway with the center less mirror (710). This further means that "cylindrical" aberration (the cylindrical equivalent of spherical aberration) is of lesser concern so that, in many cases, a simple cylindrical lens will likely suffice, and an acylindric (the cylindrical equivalent of aspheric) lens will not be needed. In order to cause this desired splitting into two beams by the first lens (713) one can
10 take a simple plano-convex cylindrical lens (fig. 7A, 13), and on the flat side facing the laser glue three small prisms (35, 36 and 37). In figure 7A, for clarity, the prisms (35, 36 and 37) are shown reflecting light from their external surfaces (as if they were silvered, for example), and it is obvious that it would be impractical to glue the outermost prisms (36 and 37) in that orientation. In practice, the outer
15 prisms (36 and 37) can be reoriented to have their reflections from an interior surface, and in that orientation they could easily be glued to the dispersing lens (13). The orientation of the prisms, either way, is such that the laser is split and the two halves of the laser light are redirected to the outer edges of the plano-convex dispersing lens (713) as shown in figure 7A. Here, again, "cylindrical" aberration
20 becomes of lesser concern for this lens (713) since only its edges are being used. Figure 7A shows the resulting pathways for four of the light rays (two inner and two outer) in this optical arrangement (the pathway of these rays passing through lens 714 are shown as solid lines and the pathway of these rays passing under
25 714 are not shown) and it can be seen that most of the laser light is utilized.

In this, fourth, embodiment the Digital Vibration Transducer (DVT) is used to digitize, and remove distortion from, a speaker. Since speakers have high power, sensitivity will not be the prime requirement of a DVT used in this application. The configuration as outlined in the second embodiment would, therefore, be
30 recommended. In figure 8 the linkage arm (816) of the DVT (shown as the enclosed unit, 43) is attached to the rear of the speaker cone (39) while the

enclosure of the DVT (43) is firmly attached to the speaker frame (40). The attachment is made so that when the speaker cone (39) is at rest the rotational mirror will also be at its rest position (0 angle of deviation). At this point there are a number of different ways to proceed. The DVT's output could be digitally compared with the digital signal of the source device (44), if that device were a CD player, for example, and the reader could easily think of other configurations, however, only the following configuration will be described since it is the most practical and versatile for the arrangement of most existing sound systems. In this configuration, the output of the DVT's binary counter (U20) is first converted to analog by a DAC (Digital to Analog Converter), U30, and this analog signal is compared by comparator U31 with the analog signal coming from the pre-amp (41) of the sound system. Any discrepancy between the pre-amp signal and the DVT's analog signal from the speaker causes a correction signal. This correction signal now becomes the input to the power amp (42) in place of the pre-amp signal that was originally connected to the power amp input. Provided that the power amp (42) has good damping and phase reproduction, the result of this configuration is that the speaker excursions will, virtually, perfectly match the signal leaving the system's pre-amp and any distortions introduced after, by the speaker, for example, will be removed by the feedback loop. In this configuration it may be desirable to remotely locate the DVT's counter (U20) with the DAC (U30) and comparator (U31) so that only a coaxial cable need run from the speaker (carrying waveform G), along with one line to power the DVT (if it is not using power from the power amp signal accessible within the speaker), and one line to carry the direction signal (H) to the counter (U20). Again, while other configurations for this DVT feedback system are possible, the intended end result for all of them is the same, removal of distortion from the speaker.

TECHNICAL DISCUSSION

Frequency Response. While the Digital Vibration Transducer (DVT) has no lower frequency limit (this is because the encoding sensor will output pulses no matter how slowly the laser sweeps across it), the upper frequency limit is almost entirely dependent on the mass of the dynamic components. As illustrated in the third embodiment that mass can be made very low. How low that mass can become is technology dependent and can not be fixed absolutely. It is reasonable to expect that, with present technology, the device can achieve a better than 100 kHz response. The frequency range is dependent on the dynamic range of the device. This is because signals of different frequencies but the same power can vary by orders of magnitude in the amount of diaphragm excursion, and therefore, the amount of laser deflection across the encoding plate, they cause. Thus a 20 kHz signal will produce one millionth of the laser deflection that a 20 Hz signal will produce at the same power. If both these signals are within the dynamic range of the device (i.e. 1,048,560 fold for a 20 bit device) the response will be flat to within +/- 3 dB, at that power level, and the greatest deviation from absolutely flat will be at the upper frequency when the least number of bits are being invoked. Thus, for a 20 bit device intended to start responding at 20 Hz, the response curve will be flat until the graph approaches 20 kHz where it will curve exponentially to +/- 3 dB at 20 kHz. The response curve for a 22 bit device will have the same shape, but with the range going from 20 Hz to 80 kHz, and for a 24 bit device that curve will have a range from 20 Hz to 320 kHz. These response curves are characteristic of all digital devices responding to a flat signal source, and they are in contrast to those one would get from analog devices, particularly of the moving coil type. Analog moving coil detectors such as dynamic microphones and magnetic phono cartridges are "velocity sensitive" and for a given coil excursion will produce a higher output voltage the higher the frequency. As a result such devices tend to perform more poorly at low frequencies and suffer low frequency limitations.

The limitations of the DVT, if it is receiving its vibrations from a perfectly suspended membrane, is that a large part of its dynamic range is being used to capture the extreme variation of excursion caused by the top and bottom frequencies (when they have equivalent power), and this leaves less dynamic range left over to capture variations in power throughout that frequency range. This can be solved by using a sound sensing membrane that is not perfectly suspended but is, rather, considerably non linear. Since the non linearity can be removed from the output (by digital calibration) distortions from such a membrane will not be a concern and, instead, advantage can be taken from the dynamic compressive effects of a membrane so suspended. That is, a membrane suspension can be used that has little resistance to small excursions but has considerable resistance to large excursions, and thus, amplitude compression takes place at the sound sensing component itself. This then, after decompression by the calibration circuitry, results in a signal having a several fold increase in dynamic range over that available from the encoding sensor itself. This configuration would be optimal when the application is for detecting the wide frequency range that occurs with music.

When the DVT is used to digitize a speaker, one is invariably dealing with speakers dedicated to a narrower range of the musical frequency spectrum (i.e. woofers, tweeters etc.), and since separate DVTs will be attached to each speaker, the same full spectrum concern of the above application does not apply. Similarly, when the application is primarily concerned with detecting large amplitude variations (seismology, for example) one generally finds the frequency range of such signals to be much narrower than that demanded by music. Ultimately, then, it is found that the most fundamental parameters, which set all the other performance characteristics and limitations for the DVT, are sensitivity and dynamic range.

Sensitivity. High sensitivity is a quality desired when the DVT is used for detecting weak signals such as in the application of a sensitive microphone or

hydrophone. The three factors most influencing the sensitivity of the digital vibration detector are:

- 5 the mass of the mirror component with the linkage arm and its bearings,
- the opto-mechanical gain of the levering arm,
- the effective pitch of the encoder and the laser wavelength on which the pitch depends.

10 The effective pitch of the encoder depends first on the actual pitch of a single encoder strip and this is limited to about 0.5 microns if a blue (440 nm) solid state laser is used. The effective pitch then depends on the maximum number of encoder strips that can be in the stack, and while the maximum possible number has yet to be established, a stack of 100 levels, giving an effective pitch of 5 nm, with a blue laser, appears to be within the limits of feasibility.

15 The opto-mechanical gain of the arrangement depends on the distance between the linkage bearing axis and the mirror's rotational axis as compared with the distance from the mirror axis to the laser focus point. While this relationship might appear to allow for very large values of gain, in practice, it is desirable to keep the
20 opto-mechanical gain as low as practical since its value also multiplies the mass of the mirror in terms of rotational inertia (due to the dynamics of a class three lever) and so, any increasing of opto-mechanical gain would be of limited value if not accompanied by a corresponding reduction in the mass of the mirror. For this discussion a gain of about 50 seems practical for use with a very light mirror
25 arranged as in the third embodiment, when the objective is for optimum sensitivity.

30 While it is difficult to establish an absolute limit to how low the mass of the mirror assembly and linkage can be made, measurements show that, in the configuration of the third embodiment, and using an opto-mechanical gain of 50, an inertia of less than 0.1 grams can be readily achieved, and this even with including the mass of a diaphragm (without the weight of a copper coil a diaphragm can be

extremely light). Such an inertia in combination with an effective encoder pitch of 5 nm, and the 50 times opto-mechanical gain suggested (giving a resolution to 0.1 nm of movement along the linkage arm), would be sensitive to a force of 0.00002 dynes-cm². Using a 1 cm² diaphragm this would correspond to a sensitivity of -20 dB or lower, depending on the frequency. While this is an extremely high sensitivity, it can be higher still by just using a larger diaphragm. 0 dB is generally considered to be the limit of detectability for human hearing and microphones.

Dynamic range. The dynamic range is dependent of the number of bits utilized or quantization level, so that for 20 bits the range is 120 dB, for 22 bits the range is 132 dB and for 24 bits the range is 144 dB. The maximum number of bits possible is dependent on the effective encoder pitch and the length of the optical lever arm, that is, the smaller the effective pitch the larger the number of pickets that can fit into the arc along which the mirror can sweep the laser line, and the longer the lever arm (radius) the longer will be the arc of the sweep. In the example of the third embodiment, a quantization level of at least 24 bits can be easily achieved if the encoder stack consists of enough levels. Further, the overall dynamic range is extended if a non linear, dynamically compressive, diaphragm is used, or if a non linear resistance is introduced anywhere among the dynamic parts of the DVT. This, however, would not increase the instantaneous dynamic range which will remain around the level available from the encoding sensor itself.

Distortion. As outlined in the background the distortion can be brought to exceptionally low values, well below 1%, and particularly when calibration circuitry is applied to the output of the binary counter to compensate for any non linearities in the system. If this circuitry is, further, designed to operate in real time there will be no sampling rate, and thus, sampling distortions will not be introduced. That is, since a CD samples at 44.1 kHz, for example, the original analog signal has to be chopped not only vertically (in amplitude) by quantization, but also horizontally (in time) by sampling. So a 5 kHz signal on a CD, for example, will be chopped about 9 times, and thus, the vertical wave shape of one such cycle will be approximated

by no more than 9 discrete levels of amplitude. This constitutes a considerable amount of distortion to which the real time output of the DVT will not be subject. To achieve this real time (not counting the inconsequential < 1 microsecond circuit delay) transducing, meaning that the DVT's output changes every moment the laser passes a picket, it is necessary for the DVT's binary counter to be able to count at very high frequencies (VHF), and in the most demanding of applications, perhaps at ultra high frequencies (UHF). Thus, if a sine wave at 10 Hz were to cause near maximum excursion on a 24 bit DVT the pulses to the counter would be arriving at about 260 MHz. Since higher signal frequencies cause exponentially smaller excursions the fastest pulse rate will occur at the lowest signal frequencies, and so for a 300 MHz counter (well within today's technology) real time instantaneous "sampling", requiring no clock, is achieved. The end result is that all the types of distortion figures for the DVT will be orders of magnitude better than those of prior transducer art.

Noise. The signal to noise ratio (s/n) of a digital device is generally considered as equal to its dynamic range. This is to allow for the possibility of there being an error of the smallest bit. For example, if the laser line were at rest, and on the exact edge of a picket so that the sensor behind was on the exact threshold of either on or off, even thermal variation in the sensor could be the deciding factor between a one or a zero. The dynamic range, however, is so great that, for all intents and purposes, noise can be considered as nonexistent, and is, at any rate, far superior to any analog counterpart.

Gain. The gain of the DVT is essentially a function of the opto-mechanical gain of the levering arrangement. This can be made adjustable by the introduction of a means to make the distance between the linkage bearing axis and the mirror's rotational axis variable. Also, the addition of another lever intervening between the linkage arm and the vibration source, and having an adjustable axis point, would allow for adjustable gain. Both of these methods, however, require the addition of components to the dynamic portion of the DVT, and their added weight will

increase the inertia of the system, thereby reducing sensitivity. When sensitivity is not the prime concern, such as with vocal microphones detecting well above 0 dB, such adjustability options are acceptable and often desirable. When the utmost sensitivity is required, such as with a hydrophone, a 24 bit non adjustable version would be required. Since such a DVT can have a range of -20 to 124 dB there should be no problem. If a wider range of loudness is expected then two DVTs can be used to cover that range, such as one covering -30 to 114 dB and one covering 30 to 174 dB.

Resolution. The resolution is the smallest amount of movement detectable and is found by dividing the effective pitch by the opto-mechanical gain. When sensitivity and size is not the prime concern, very high resolution (below 0.1 nm) is easily possible as it is simply a matter of increasing the opto-mechanical gain. If the vibration source has considerable power but extremely small motion the DVT needs only to be configured having a high opto-mechanical gain and the smallest effective encoder pitch as is practical. If a 100 level encoder stack is used with a blue laser to give 5 nm effective pitch it is necessary for the laser to have enough power be spread over the 100 levels, and still be bright enough on each individual encoding strip to effectively illuminate it. Such lasers, with their heat sinks, can have sizes approaching that of an AA cell battery. The opto-mechanical gain can be increased many fold by adding another lever intervening between the linkage arm and the vibration source, further adding some additional size to the DVT. Thus, the main collateral effects of increasing resolution are an increase in size and power consumption.

Size and weight. Small size and weight are mostly needed in the application of hand held or headset mounted microphones. Since these are vocal microphones, and the source is very close, high resolution and sensitivity are not the primary objectives. Thus, a 20 bit DVT constructed to have its dynamic range (at the encoding sensor) from 10 to 130 dB, and using a non linear suspension to extend that range to about 150 dB, would be the optimum choice. By using a 50 level

encoder stack, and a small red laser, the effective encoder pitch would be 20 nm, and this would require an encoding sensor having a 2.1 cm long arc. Using this with a mirror having a 2.5 cm radius to the focal point, locating the counting and calibrating circuitry remotely, and by folding the optical pathways with static mirrors so as to economize the use of space, the transducing component could be expected to fit inside a cylinder 4 cm long by 2.5 cm wide, or somewhat shorter than a C cell battery. Further, if composite materials and acrylic optics are used, the weight of this example could be expected to be under one or two ounces.

Robustness and longevity. While some of the components of the DVT have the delicacy of watch works, this actually serves to make the device as robust as a watch. That is, since the rotating mirror can weigh less than 0.01 g, it only requires a bearing pressure of one gram to prevent the mirror from dislodging under a shock of more than 100 G force. Further, with acrylic optical parts, solid state and CD type structures comprising the remainder of the DVT, the durability of the device, properly enclosed, can easily match that of analog microphones designed for the most demanding environments. The only component of the DVT prone to significant lifetime limitations is the laser. Blue and red solid state diode lasers operating at maximum rated power have lifetime expectancies of over 10,000 hours, or over 13 months of continuous use. When under driven, though, red lasers can surpass 100,000 hours before failure, or over 11 years continuous use. The figure for blue diode lasers is unknown due to their recent development. In any event, the DVT can be designed to use a standardized laser element that is easily replaced.

Ease of manufacturing. The construction of the DVT is typical of the micro fabrication techniques used for VLSI (very large scale integration) chips, CD pressing and precision acrylic molding. A molded framework can be used to which all the components are easily attached leaving only the need for several precision alignments, as with any premium microphone. The mounting of the bearings to the their components will be critical, but no more so than the attachment of a precision

copper coil to a diaphragm and aligning it to a magnet, for example. Thus, the DVT can be considered as lending itself to mass production as much do analog microphones.

- 5 **Cost.** The expense of a copper coil and a high quality magnet are not among those incurred in the manufacture of the DVT. Instead, expense will be concentrated in the bearings, optical components, and perhaps the laser, given that cost of the electronic sections, when formed as VLSI chips, drop dramatically with mass production. The overall cost of the DVT can then be expected to be
10 comparable to the best of its analog counterparts and, at least, commensurate with the exceptional performance of the device.

In addition to the uses already outlined, the DVT embodies the trend towards digital finally brought to the audio transducer, and a direct digital interfacing to the
15 outside analog world. Thus, since environmental vibrations are analog by nature, the digital encoding sensor in the DVT is essentially an analog to digital converter (ADC), converting the analog motion of the levered laser directly into an electronic digital signal. This markedly differs from a system where, say, a conventional electronic ADC converts the analog electrical signal from a dynamic microphone's
20 sensing coil. Such a system does not lend itself to subsequent calibration to remove non-linearities and, further, there will always be an electrical noise component, prior to the conversion to digital, which, further still, will always be subject to thermal variations. Since the optical conversion to digital of the DVT avoids all of these and other limitations the device introduces an entirely new level
25 of possibilities in transducer technology.

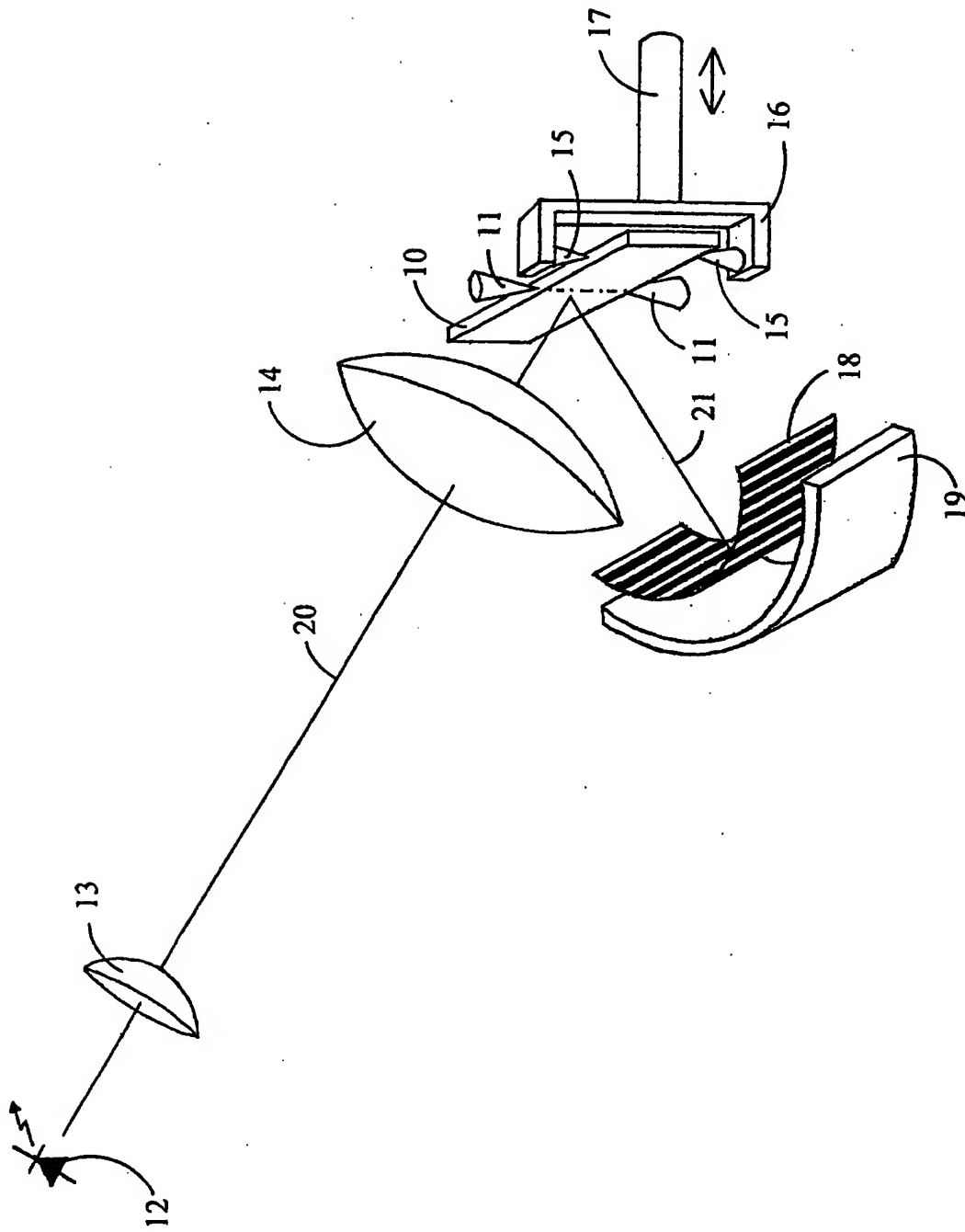
The full implications of the DVT access to this new landscape in signal detection can only be guessed at, but what is clear is that the present invention provides an
30 unparalleled access to signal manipulation that has distinct and dramatic advantages over prior technology. The nature of the signals provided by the DVT

are of such a unique characteristic that only extensive applications will illuminate the full extent of the utilities and qualities afforded by the present invention.

5 Although the present invention has been explained hereinabove by way of a preferred embodiment thereof, it should be pointed out that any modifications to this preferred embodiment within the scope of the appended claims is not deemed to alter or change the nature and scope of the present invention.

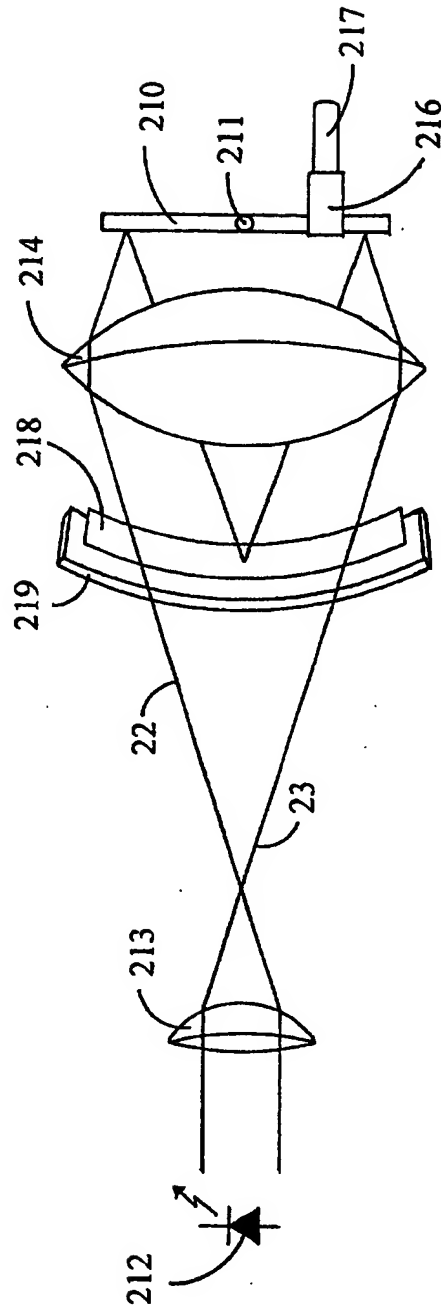
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Fig. 1



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Fig. 2



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Fig. 3A

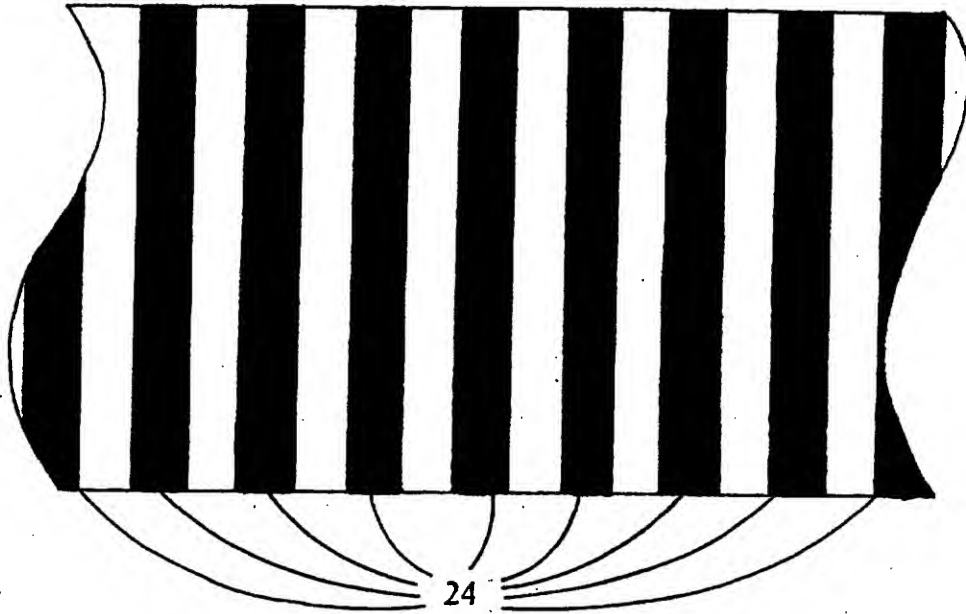
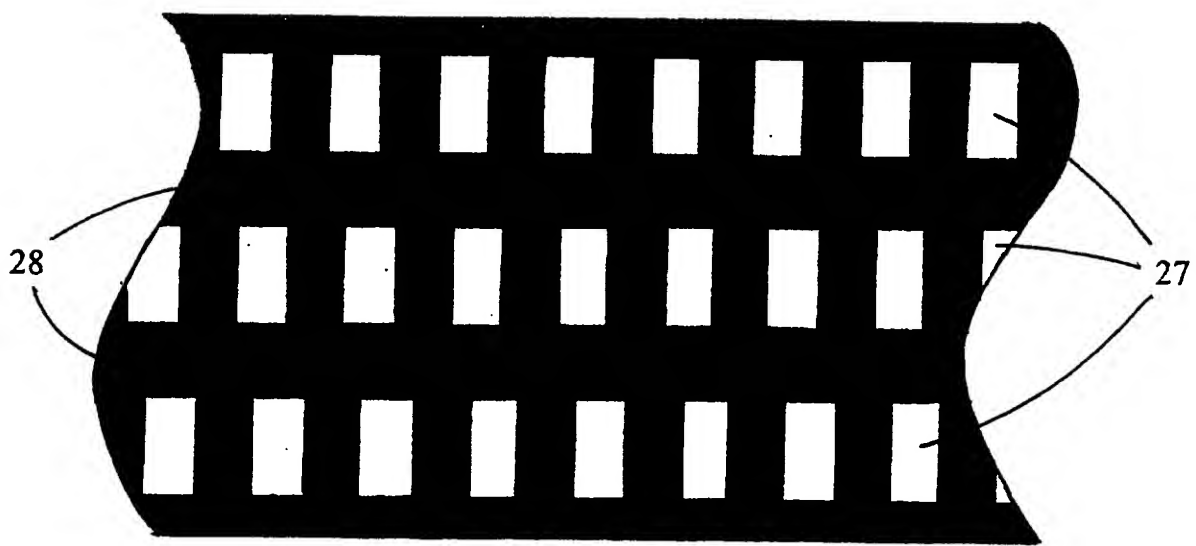


Fig. 3B



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Fig. 4A

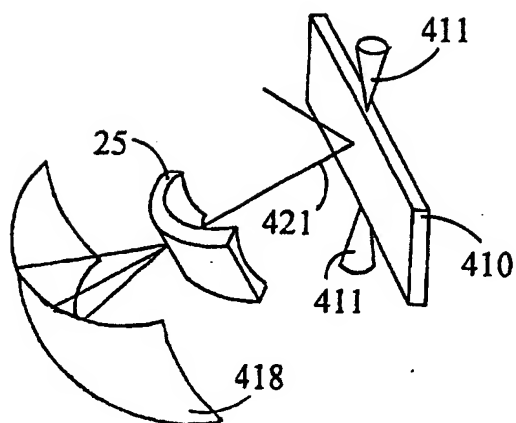
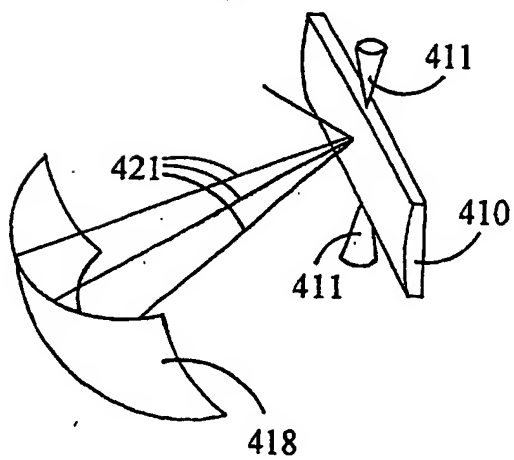
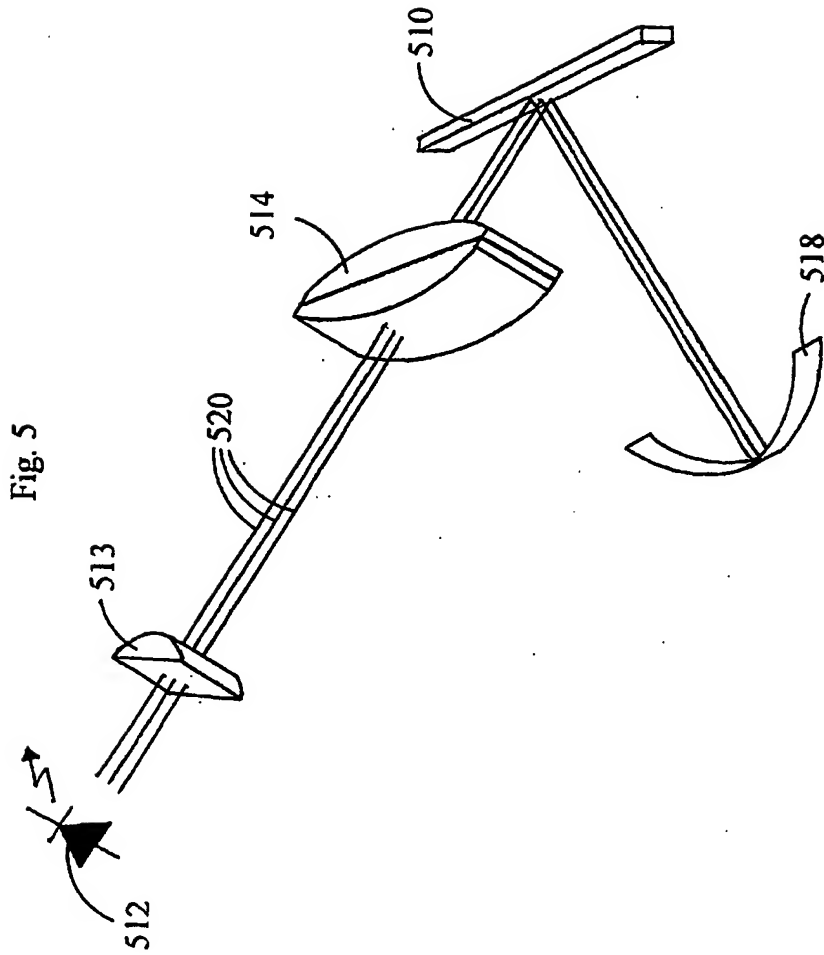


Fig. 4B



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FIGURE 6A

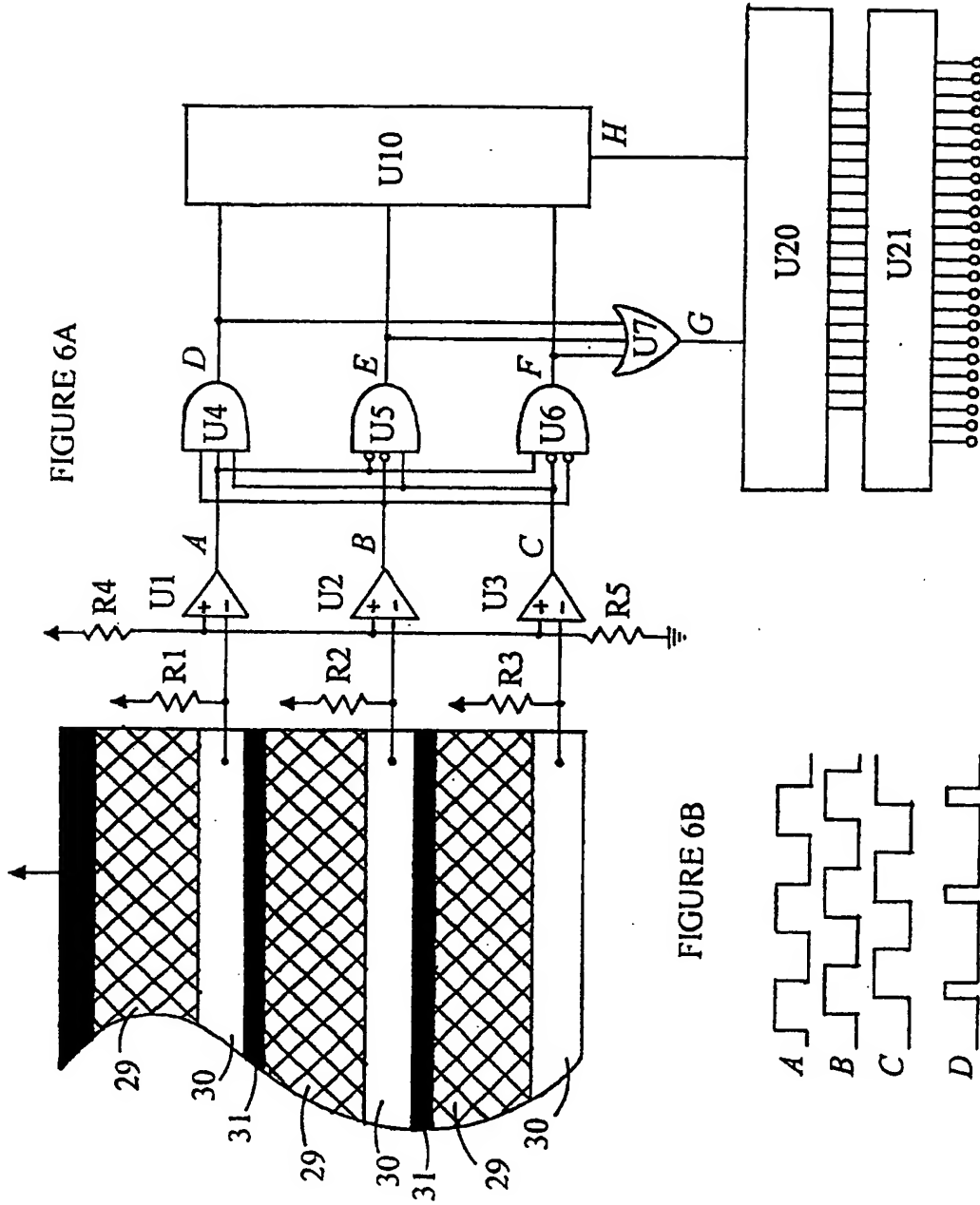
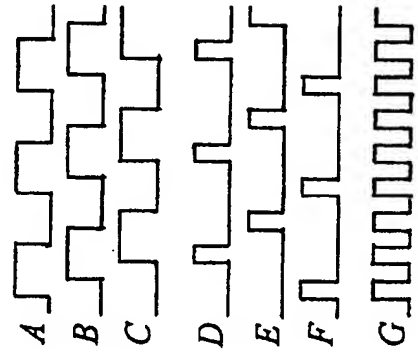


FIGURE 6B



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Fig. 7A

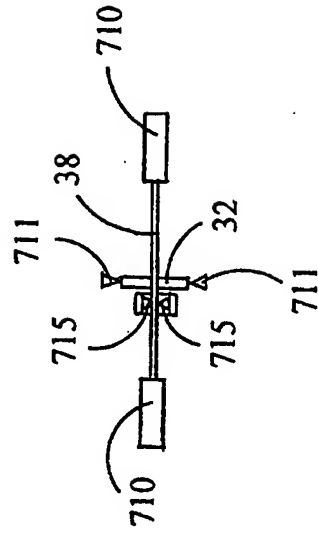
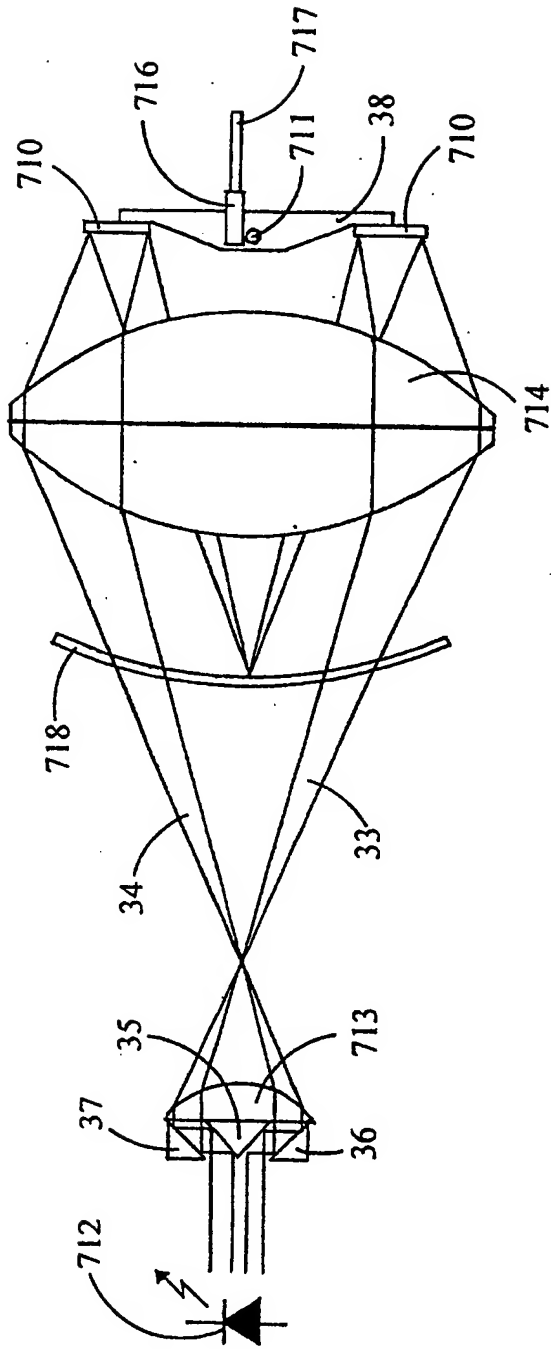


Fig. 7B

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Fig. 8

